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Patent and Trademark Office: U.S. DEPARTMENT OF COMMERCE

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UTILITY PATENT APPLICATION TRANSMITTAL

(Only for new non-provisional applications under 37 C.F.R. § 1.53(B))

Attorney Docket No.	M61.12-0316
First Inventor or Application Identifier	Xuedong Huang et al.
Title	METHOD AND APPARATUS USING SPECTRAL ADDITION FOR SPEAKER RECOGNITION
Express Mail Label No.	EL636048810US

Address To: Assistant Commissioner for Patents
Box Patent Application
Washington, DC 20231

APPLICATION ELEMENTS

See MPEP chapter 600 concerning utility patent application contents.

1. ☒ *Fee Transmittal Form *e.g., PTO/SB17)* *(Submit an original and a duplicate for fee processing)*
2. ☒ Specification *[Total Sheets* **49** *]*
   (preferred arrangement set forth below
   - Descriptive title of the Invention)
   - Cross References to Related Applications
   - Statement Regarding Fed sponsored R & D
   - Reference to Microfiche Appendix
   - Background of the Invention
   - Brief Summary of the Invention
   - Brief Description of the Drawings *(if filed)*
   - Detailed Description
   - Claim(s)
   - Abstract of the Disclosure
3. ☒ Drawing(s) *(35 U.S.C. § 113)* *[Total Sheets* **11** *]*
4. Oath or Declaration *[Total Sheets* **3** *]*
   ☒ Newly executed (original or copy)
   a. ☐ Copy from a prior application (37 C.F.R. § 1.63(d)) *(for continuation/divisional with Box 16 completed)*
   i. ☐ DELETION OF INVENTOR(S) Signed statement attached deleting inventor(s) named in the prior application, see 37 C.F.R §§1.63(d)(2) and 1.33(b).

NOTE FOR ITEMS 1 & 13: IN ORDER TO BE ENTITLED TO PAY SMALL ENTITY FEES, A SMALL ENTITY STATEMENT IS REQUIRED (37 C.F.R. § 1.27), EXCEPT IF ONE FILED IN A PRIOR APPLICATION IS RELIED UPON (37 C.F.R. § 1.28).

5. ☐ Microfiche Computer Program *(Appendix)*
6. ☐ Nucleotide and/or Amino Acid Sequence Submission *(If applicable, all necessary)*
   a. ☐ Computer Readable Copy
   b. ☐ Paper Copy (Identical to computer copy)
   c. ☐ Statement verifying identity of above copies

ACCOMPANYING APPLICATION PARTS

7. ☒ Assignment Papers (cover sheet & document(s))
8. ☐ 37 C.F.R. § 3.73(b) Statement *(when there is an assignee)* ☒ Power of Attorney
9. ☐ English Translation Document
10. ☐ Information Disclosure Statement (IDS/PTO – PTO) ☐ Copies of IDS
11. ☐ Preliminary Amendment
12. ☒ Return Receipt Postcard (MPEP 503)
13. ☐ *Small Entity Statement(s) *(PTO/SB/09-12)* ☐ Statement filed in prior application. Status still proper and desired
14. ☐ Certified Copy of Priority Document(s) *(if foreign priority is claimed)*
15. ☒ Other: Check in the amount of $952.00
    Check in the amount of $40.00

16. **If a CONTINUING APPLICATION,** *check appropriate box, and supply the requisite information below and in a preliminary amendment:*
    ☐ Continuation ☐ Divisional ☐ Continuation –in part (CIP) of prior application No: _____
    Prior application information: *Examiner* _____ *Group/Art Unit.* ___
    FOR CONTINUATION or DIVISIONAL APPS only: The entire disclosure of the prior application, from which an oath or declaration is supplied under Box 4b, is considered a part of the disclosure of the accompanying continuation or divisional application and is hereby incorporated by reference. The incorporation can only be relied upon when a portion has been inadvertently omitted from the submitted application parts.

17. CORRESPONDENCE

☐ *Customer Number or Bar Code Label* *(Insert Customer No. or Attach bar code label here)* *or* ☒ ***Correspondence address below***

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Signature	Theodore M Magee	Date	10/10/00

09685534 10/10/00

FEE TRANSMITTAL

Complete if Known	
Application No.	
Filing Date	
First Named Inventor	Xuedong Huang
Title	METHOD AND APPARATUS USING SPECTRAL ADDITION FOR SPEAKER RECOGNITION
Group Art Unit	
Examiner Name	
Atty. Docket Number	M61.12-0316

Total Amount of Payment $ **952**

METHOD OF PAYMENT (Check One)

1. ☒ The Director is hereby authorized to charge any additional fee required under 37 C.F.R. § 1.16 and 1.17, including any petition fee, and credit any over payments to Deposit Account No. 23-1123. Westman, Champlin & Kelly, P.A.

2. ☒ Check Enclosed

FEE CALCULATION

1. **BASIC FILING FEE**

Large Entity Fee Code	Large Entity Fee ($)	Small Entity Fee Code	Small Entity Fee ($)	Fee Description
101	710	201	355	☒ Utility Filing Fee
106	320	206	160	☐ Design Filing Fee
108	710	208	355	☐ Reissue Filing Fee
114	150	214	75	☐ Prov. Filing Fee

Subtotal (1) $ 710

2. **EXTRA CLAIM FEES**

	Number Claims	Prior**	Extra	Fee from Below	Fee Paid
Total	**29**	**20**	**9**	**18**	**162**
Indep.	**4**	**3**	**1**	**80**	**80**

Multiple Dependent Claims

** Insert 3 and 20, or number previously paid if greater, Reissue see below

Large Entity Fee Code	Large Entity Fee ($)	Small Entity Fee Code	Small Entity Fee ($)	Description
103	18	203	9	Claims in excess of 20
102	80	202	40	Independent claims in excess of 3
104	270	204	135	Multiple Dependent Claims
109	80	209	40	Reissue Independent Claims over Original Patent
110	18	210	9	Reissue claims in excess of 20 and over original patent

Subtotal (2) $ 242

FEE CALCULATION (Continued)

3. **ADDITIONAL FEES**

Large Entity Fee Code	Large Entity Fee ($)	Small Entity Fee Code	Small Entity Fee ($)	Fee Description	Fee Paid
105	130	205	65	Surcharge - Late filing fee or oath	
127	50	227	25	Surcharge - Late provisional Filing Fee or cover sheet	
139	130	139	130	Non-English specification	
147	2,520	147	2,520	For Filing a Request for Reexamination	
115	110	215	55	Extension for reply within first month	
116	390	216	195	Extension for reply within second month	
117	890	217	445	Extension for reply within third month	
118	1,390	218	695	Extension for reply within fourth month	
128	1,890	280	945	Extension for reply within fifth month	
120	310	220	155	Filing a brief in support of an appeal	
121	270	221	135	Request for oral hearing	
148	110	248	55	Terminal Disclaimer Fee	
140	110	240	55	Petition to Revive - unavoidable	
141	1,240	241	620	Petition to Revive - unintentional	
142	1,270	242	650	Utility/Reissue issue fee (inc. advance copies)	
143	470	243	250	Design issue fee (inc. advance copies)	
122	130	122	130	Petitions to the Commissioner	
123	50	123	50	Petitions related to provisional applications	
126	240	126	240	Submission of Information Disclosure Statement	
581	40	581	40	Recording each patent assignment per property (times number of properties)	

Other Fee (specify) ____

Subtotal (3) $

Signature Theodore M Mg
(**Theodore M. Magee**)

Reg. No. **39,758**

Date _____

Deposit Account No. 23-1123

FEE TRANSMITTAL

Complete if Known	
Application No.	
Filing Date	
First Named Inventor	Xuedong Huang
Title	METHOD AND APPARATUS USING SPECTRAL ADDITION FOR SPEAKER RECOGNITION
Group Art Unit	
Examiner Name	
Atty. Docket Number	M61.12-0316

Total Amount of Payment $ 952

METHOD OF PAYMENT (Check One)

1. ☒ The Director is hereby authorized to charge any additional fee required under 37 C.F.R. § 1.16 and 1.17, including any petition fee, and credit any over payments to Deposit Account No. 23-1123. Westman, Champlin & Kelly, P.A.

2. ☒ Check Enclosed

FEE CALCULATION

1. **BASIC FILING FEE**

Large Entity Fee Code	Large Entity Fee ($)	Small Entity Fee Code	Small Entity Fee ($)	Fee Description
101	710	201	355	☒ Utility Filing Fee
106	320	206	160	☐ Design Filing Fee
108	710	208	355	☐ Reissue Filing Fee
114	150	214	75	☐ Prov. Filing Fee

Subtotal (1) $ 710

2. **EXTRA CLAIM FEES**

	Number Claims	Prior**	Extra	Fee from Below	Fee Paid
Total	29	20	9	18	162
Indep.	4	3	1	80	80

Multiple Dependent Claims

** Insert 3 and 20, or number previously paid if greater. Reissue see below

Large Entity Fee Code	Large Entity Fee ($)	Small Entity Fee Code	Small Entity Fee ($)	Description
103	18	203	9	Claims in excess of 20
102	80	202	40	Independent claims in excess of 3
104	270	204	135	Multiple Dependent Claims
109	80	209	40	Reissue Independent Claims over Original Patent
110	18	210	9	Reissue claims in excess of 20 and over original patent

Subtotal (2) $ 242

FEE CALCULATION (Continued)

3. **ADDITIONAL FEES**

Large Entity Fee Code	Large Entity Fee ($)	Small Entity Fee Code	Small Entity Fee ($)	Fee Description	Fee Paid
105	130	205	65	Surcharge - Late filing fee or oath	
127	50	227	25	Surcharge - Late provisional Filing Fee or cover sheet	
139	130	139	130	Non-English specification	
147	2,520	147	2,520	For Filing a Request for Reexamination	
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117	890	217	445	Extension for reply within third month	
118	1,390	218	695	Extension for reply within fourth month	
128	1,890	280	945	Extension for reply within fifth month	
120	310	220	155	Filing a brief in support of an appeal	
121	270	221	135	Request for oral hearing	
148	110	248	55	Terminal Disclaimer Fee	
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123	50	123	50	Petitions related to provisional applications	
126	240	126	240	Submission of Information Disclosure Statement	
581	40	581	40	Recording each patent assignment per property (times number of properties)	

Other Fee (specify) ____

Subtotal (3) $

Signature Theodore M Mg
(Theodore M. Magee)

Reg. No. 39,758

Date _____

Deposit Account No. 23-1123

Express Mail No. EL636048810US

PATENT APPLICATION OF

XUEDONG HUANG AND MICHAEL D. PLUMPE

ENTITLED

METHOD AND APPARATUS USING SPECTRAL ADDITION FOR SPEAKER RECOGNITION

3065P-1360US

Docket No. M61.12-0316

METHOD AND APPARATUS USING SPECTRAL ADDITION FOR SPEAKER RECOGNITION

BACKGROUND OF THE INVENTION

The present invention relates to speaker recognition. In particular, the present invention relates to training and using models for speaker recognition.

A speaker recognition system identifies a person from their speech. Such systems can be used to control access to areas or computer systems as well as tailoring computer settings for a particular person.

In many speaker recognition systems, the system asks the user to repeat a phrase that will be used for recognition. The speech signal that is generated while the user is repeating the phrase is then used to train a model. When a user later wants to be identified by their speech, they repeat the identification phrase. The resulting speech signal, sometimes referred to as a test signal, is then applied against the model to generate a probability that the test signal was generated by the same person who produced the training signals.

The generated probability can then be compared to other probabilities that are generated by applying the test signal to other models. The model that produces the highest probability is then

considered to have been produced by the same speaker who generated the test signal. In other systems, the probability is compared to a threshold probability to determine if the probability is sufficiently high to identify the person as the same person who trained the model. Another type of system would compare the probability to the probability of a general model designed to represent all speakers.

The performance of speaker recognition systems is affected by the amount and type of background noise in the test and training signals. In particular, the performance of these systems is negatively impacted when the background noise in the training signal is different from the background noise in the test signal. This is referred to as having mismatched signals, which generally provides lower accuracy than having so-called matched training and testing signals.

To overcome this problem, the prior art has attempted to match the noise in the training signal to the noise in the testing signal. Under some systems, this is done using a technique known as spectral subtraction. In spectral subtraction, the systems attempt to remove as much noise as possible from both the training signal and the test signal. To remove the noise from the training signal, the systems first collect noise samples during pauses in the speech found in the training signal. From these samples, the mean of each frequency component of the noise is determined. Each frequency mean is then

subtracted from the remaining training speech signal. A similar procedure is followed for the test signal, by determining the mean strength of the frequency components of the noise in the test signal.

Spectral subtraction is less than ideal as a noise matching technique. First, spectral subtraction does not remove all noise from the signals. As such, some noise remains mismatched. In addition, because spectral subtraction performs a subtraction, it is possible for it to generate a training signal or a test signal that has a negative strength for a particular frequency. To avoid this, many spectral subtraction techniques abandon the subtraction when the subtraction will result in negative strength, using a flooring technique instead. In those cases, the spectral subtraction technique is replaced with a technique of attenuating the particular frequency.

For these reasons, a new noise matching technique for speaker recognition is needed.

SUMMARY OF THE INVENTION

A method and apparatus for speaker recognition is provided that matches the noise in training data to the noise in testing data using spectral addition. Under spectral addition, the mean and variance for a plurality of frequency components are adjusted in the training data and the test data so that each mean and variance is matched in a resulting matched training signal and matched test signal. The adjustments made to the training data

and test data add to the mean and variance of the training data and test data instead of subtracting from the mean and variance.

BRIEF DESCRIPTION OF THE DRAWINGS

FIG. 1 is a block diagram of one computing environment in which the present invention may be practiced.

FIG. 2 is a block diagram of an alternative computing environment in which the present invention may be practiced.

FIG. 3 is a flow diagram of one embodiment of a method of speaker recognition under the present invention.

FIG. 4 is a more detailed block diagram of a speaker recognition system of one embodiment of the present invention.

FIG. 5 is a more detailed block diagram of a noise matching component under one embodiment of the present invention.

FIG. 6 is a graph of a speech signal.

FIG. 7 is a flow diagram of a method of matching variances for a frequency component under one embodiment of the present invention.

FIG. 8 is a graph of the strength of a frequency component as a function of time.

FIG. 9 is a graph of the strength of a frequency component as a function of time for a noise segment showing the mean of the strength.

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FIG. 10 is a graph of the strength of the frequency component of FIG. 9 after subtracting the mean.

FIG. 11 is a graph of the strength of the frequency component of FIG. 10 after multiplying by a gain factor.

FIG. 12 is a graph of the strength of the frequency component for a segment of the test signal or training signal.

FIG. 13 is a graph of the segment of FIG. 12 after subtracting the variance pattern of FIG. 11.

FIG. 14 is a graph of the segment of FIG. 13 after adding the absolute value of the most negative value to all values of the frequency component.

FIG. 15 is a flow diagram of a method of matching means under one embodiment of the present invention.

FIG. 16 is a graph of the strength of a frequency component of one of the test signal or training signal.

FIG. 17 shows the graph of FIG. 16 after adding the difference in means.

DETAILED DESCRIPTION OF ILLUSTRATIVE EMBODIMENTS

FIG. 1 illustrates an example of a suitable computing system environment 100 on which the invention may be implemented. The computing system environment 100 is only one example of a suitable computing environment and is not intended to suggest

any limitation as to the scope of use or functionality of the invention. Neither should the computing environment 100 be interpreted as having any dependency or requirement relating to any one or combination of components illustrated in the exemplary operating environment 100.

The invention is operational with numerous other general purpose or special purpose computing system environments or configurations. Examples of well known computing systems, environments, and/or configurations that may be suitable for use with the invention include, but are not limited to, personal computers, server computers, hand-held or laptop devices, multiprocessor systems, mircoprocessor-based systems, set top boxes, programmable consumer electronics, network PCs, minicomputers, mainframe computers, distributed computing environments that include any of the above systems or devices, and the like. In addition, the invention may be used in a telephony system.

The invention may be described in the general context of computer-executable instructions, such as program modules, being executed by a computer. Generally, program modules include routines, programs, objects, components, data structures, etc. that perform particular tasks or implement particular abstract data types. The invention may also be practiced in distributed computing environments where tasks are performed by remote processing devices that are linked through a

communications network. In a distributed computing environment, program modules may be located in both local and remote computer storage media including memory storage devices.

With reference to FIG. 1, an exemplary system for implementing the invention includes a general-purpose computing device in the form of a computer 110. Components of computer 110 may include, but are not limited to, a processing unit 120, a system memory 130, and a system bus 121 that couples various system components including the system memory to the processing unit 120. The system bus 121 may be any of several types of bus structures including a memory bus or memory controller, a peripheral bus, and a local bus using any of a variety of bus architectures. By way of example, and not limitation, such architectures include Industry Standard Architecture (ISA) bus, Micro Channel Architecture (MCA) bus, Enhanced ISA (EISA) bus, Video Electronics Standards Association (VESA) local bus, and Peripheral Component Interconnect (PCI) bus also known as Mezzanine bus.

Computer 110 typically includes a variety of computer readable media. Computer readable media can be any available media that can be accessed by computer 110 and includes both volatile and nonvolatile media, removable and non-removable media. By way of example, and not limitation, computer readable media may comprise computer storage media and communication media. Computer storage media

includes both volatile and nonvolatile, removable and non-removable media implemented in any method or technology for storage of information such as computer readable instructions, data structures, program modules or other data. Computer storage media includes, but is not limited to, RAM, ROM, EEPROM, flash memory or other memory technology, CD-ROM, digital versatile disks (DVD) or other optical disk storage, magnetic cassettes, magnetic tape, magnetic disk storage or other magnetic storage devices, or any other medium which can be used to store the desired information and which can be accessed by computer 100. Communication media typically embodies computer readable instructions, data structures, program modules or other data in a modulated data signal such as a carrier wave or other transport mechanism and includes any information delivery media. The term "modulated data signal" means a signal that has one or more of its characteristics set or changed in such a manner as to encode information in the signal. By way of example, and not limitation, communication media includes wired media such as a wired network or direct-wired connection, and wireless media such as acoustic, FR, infrared and other wireless media. Combinations of any of the above should also be included within the scope of computer readable media.

The system memory 130 includes computer storage media in the form of volatile and/or nonvolatile memory such as read only memory (ROM) 131

and random access memory (RAM) 132. A basic input/output system 133 (BIOS), containing the basic routines that help to transfer information between elements within computer 110, such as during start-up, is typically stored in ROM 131. RAM 132 typically contains data and/or program modules that are immediately accessible to and/or presently being operated on by processing unit 120. By way o example, and not limitation, FIG. 1 illustrates operating system 134, application programs 135, other program modules 136, and program data 137.

The computer 110 may also include other removable/non-removable volatile/nonvolatile computer storage media. By way of example only, FIG. 1 illustrates a hard disk drive 141 that reads from or writes to non-removable, nonvolatile magnetic media, a magnetic disk drive 151 that reads from or writes to a removable, nonvolatile magnetic disk 152, and an optical disk drive 155 that reads from or writes to a removable, nonvolatile optical disk 156 such as a CD ROM or other optical media. Other removable/non-removable, volatile/nonvolatile computer storage media that can be used in the exemplary operating environment include, but are not limited to, magnetic tape cassettes, flash memory cards, digital versatile disks, digital video tape, solid state RAM, solid state ROM, and the like. The hard disk drive 141 is typically connected to the system bus 121 through a non-removable memory interface such as interface 140, and magnetic disk drive 151 and optical disk drive

155 are typically connected to the system bus 121 by a removable memory interface, such as interface 150.

The drives and their associated computer storage media discussed above and illustrated in FIG. 1, provide storage of computer readable instructions, data structures, program modules and other data for the computer 110. In FIG. 1, for example, hard disk drive 141 is illustrated as storing operating system 144, application programs 145, other program modules 146, and program data 147. Note that these components can either be the same as or different from operating system 134, application programs 135, other program modules 136, and program data 137. Operating system 144, application programs 145, other program modules 146, and program data 147 are given different numbers here to illustrate that, at a minimum, they are different copies.

A user may enter commands and information into the computer 110 through input devices such as a keyboard 162, a microphone 163, and a pointing device 161, such as a mouse, trackball or touch pad. Other input devices (not shown) may include a joystick, game pad, satellite dish, scanner, or the like. These and other input devices are often connected to the processing unit 120 through a user input interface 160 that is coupled to the system bus, but may be connected by other interface and bus structures, such as a parallel port, game port or a universal serial bus (USB). A monitor 191 or other type of display device is also connected to the

system bus 121 via an interface, such as a video interface 190. In addition to the monitor, computers may also include other peripheral output devices such as speakers 197 and printer 196, which may be connected through an output peripheral interface 190.

The computer 110 may operate in a networked environment using logical connections to one or more remote computers, such as a remote computer 180. The remote computer 180 may be a personal computer, a hand-held device, a server, a router, a network PC, a peer device or other common network node, and typically includes many or all of the elements described above relative to the computer 110. The logical connections depicted in FIG. 1 include a local area network (LAN) 171 and a wide area network (WAN) 173, but may also include other networks. Such networking environments are commonplace in offices, enterprise-wide computer networks, intranets and the Internet.

When used in a LAN networking environment, the computer 110 is connected to the LAN 171 through a network interface or adapter 170. When used in a WAN networking environment, the computer 110 typically includes a modem 172 or other means for establishing communications over the WAN 173, such as the Internet. The modem 172, which may be internal or external, may be connected to the system bus 121 via the user input interface 160, or other appropriate mechanism. In a networked environment, program modules depicted relative to the computer

110, or portions thereof, may be stored in the remote memory storage device. By way of example, and not limitation, FIG. 1 illustrates remote application programs 185 as residing on remote computer 180. It will be appreciated that the network connections shown are exemplary and other means of establishing a communications link between the computers may be used.

FIG. 2 is a block diagram of a mobile device 200, which is an exemplary computing environment. Mobile device 200 includes a microprocessor 202, memory 204, input/output (I/O) components 206, and a communication interface 208 for communicating with remote computers or other mobile devices. In one embodiment, the afore-mentioned components are coupled for communication with one another over a suitable bus 210.

Memory 204 is implemented as non-volatile electronic memory such as random access memory (RAM) with a battery back-up module (not shown) such that information stored in memory 204 is not lost when the general power to mobile device 200 is shut down. A portion of memory 204 is preferably allocated as addressable memory for program execution, while another portion of memory 204 is preferably used for storage, such as to simulate storage on a disk drive.

Memory 204 includes an operating system 212, application programs 214 as well as an object store 216. During operation, operating system 212 is preferably executed by processor 202 from memory 204.

Operating system 212, in one preferred embodiment, is a WINDOWS® CE brand operating system commercially available from Microsoft Corporation. Operating system 212 is preferably designed for mobile devices, and implements database features that can be utilized by applications 214 through a set of exposed application programming interfaces and methods. The objects in object store 216 are maintained by applications 214 and operating system 212, at least partially in response to calls to the exposed application programming interfaces and methods.

Communication interface 208 represents numerous devices and technologies that allow mobile device 200 to send and receive information. The devices include wired and wireless modems, satellite receivers and broadcast tuners to name a few. Mobile device 200 can also be directly connected to a computer to exchange data therewith. In such cases, communication interface 208 can be an infrared transceiver or a serial or parallel communication connection, all of which are capable of transmitting streaming information.

Input/output components 206 include a variety of input devices such as a touch-sensitive screen, buttons, rollers, and a microphone as well as a variety of output devices including an audio generator, a vibrating device, and a display. The devices listed above are by way of example and need not all be present on mobile device 200. In addition, other input/output devices may be attached

to or found with mobile device 200 within the scope of the present invention.

Under the present invention, an apparatus and method are provided that improve the matching of noise between training data and test data. FIG. 3 shows one embodiment of a method for performing such matching.

In step 300 of FIG. 3, a speaker recognition system 398, shown in FIG. 4, receives and stores a training signal. The training signal is received through a microphone 404, which detects a speech signal produced by speaker 400 and additive noise 402. Typically, the training signal is generated by the speaker reading a short identification text such as "Log me in". Microphone 404 converts the acoustic signal produced by speaker 400 and additive noise 402 into an electrical analog signal that is provided to an analog-to-digital converter 406. Analog-to-digital converter 406 samples the analog signal to produce digital values that are stored in a training data storage 407. Although the training data is shown as being stored as time domain values, those skilled in the art will recognize that the training data may be stored in the frequency domain or as a collection of feature vectors.

At step 302 of FIG. 3, speaker recognition system 398 receives a test signal from speaker 400 along with different additive noise 402. This test signal is typically generated by repeating the

identification phrase that was used to generate the training data. Like the training signal, the test signal passes through a microphone and an analog-to-digital converter. Although only one microphone 404 and analog-to-digital converter 406 are shown in FIG. 4, those skilled in the art will recognize that a different microphone and analog-to-digital converter can be used during testing than was used during training. The digital values produced by analog-to-digital converter 406 are then provided to a noise matching component 408, which also receives the training data that is stored in training data storage 407.

At step 304, noise matching component 408 identifies and stores the spectrum of selected samples found in training signal and the test signal. The elements for performing this identification are shown in more detail in FIG. 5.

In FIG. 5, the test data and training data are each provided to a respective frame construction unit 500, 501, which divide the respective signals into frames, each typically 25 milliseconds long and each starting 10 milliseconds after the start of the previous frame. Each frame is multiplied by a respective window 502, 503, which is typically a Hamming window or a Hanning window. The resulting windows of speech are provided to respective noise identification units 504 and 505

Noise identification units 504 and 505 identify which frames contain only noise and which

frames contain a combination of noise and speech. As can be seen in FIG. 6, a speech signal contains active speech regions and non-active speech regions. In FIG. 6, time is shown along horizontal axis 600 and the amplitude of the speech signal is shown along vertical axis 602. The speech signal of FIG. 6 includes one active region 604 that contains both noise and speech and two non-active regions 606 and 608 that contain only noise and represent periods where the speaker has paused.

Noise identification units 504 and 505 can use any of a number of known techniques to classify the frames as speech or noise. As is known in the art, these techniques can operate on the windowed speech signal directly or on transformations of the speech signal such as Fast Fourier Transform values or mel-cepstrum features.

When noise identification units 504 and 505 of FIG. 5 classify a frame as a noise frame, they pass the noise frame through a respective Fast Fourier Transform (FFT) 506, 508. Each FFT 506, 508 converts the values in the noise frame into a collection of frequency values representing the spectrum of the signal in the frame. These frequency values represent the relative strength of each frequency component in the spectrum of the signal and can be amplitude values or energy values. The FFTs produce complex numbers for the frequency values, energy is calculated as the square of the real value added to the square of the imaginary value. The

amplitude is simply the square root of the energy. (Note that in the present application, and specifically in the claims, a reference to a strength value for a frequency component can be interpreted as either an amplitude value or an energy value.)

The strength values are stored in a noise storage 510. As is shown in FIG. 5, noise storage 510 is divided into two sections, a training storage 512, which contains noise frames from the training speech and a test storage 514, which contains noise frames from the test speech.

Once the spectrum of the noise frames for the training signal and test signal have been stored at step 304 of FIG. 3, the process of FIG. 3 continues at step 306. In step 306, the means and variances of a plurality of frequency components in the noise of the training signal and in the noise of the test signal are adjusted so that the means and variances are the same in both signals. This is performed by a spectral adder 516, which accesses the noise segments stored in noise storage 510. The technique for adjusting the means and variances is discussed further below in connection with FIG. 7.

Once the variances and the means of each frequency component have been matched, the matched training signal is output by spectral adder 516 to a feature extractor 410 of FIG. 4. Feature extractor 410 extracts one or more features from the training signal. Examples of possible feature extraction modules that can be used under the present invention

include modules for performing linear predictive coding (LPC), LPC direct cepstrum, perceptive linear prediction (PLP), auditory model feature extraction, and Mel-frequency cepstrum coefficients feature extraction. Note that the invention is not limited to these feature extraction modules and that other modules may be used within the context of the present invention.

Using the features extracted by feature extractor 410, the method of FIG. 3 continues at step 310 by training a model based on the features extracted from the matched training signal. This training is performed by a trainer 424 of FIG. 4 based on the training identification phrase 426.

Using the extracted features and the training phrase, trainer 424 builds an acoustic model 418. In one embodiment, acoustic model 418 is a Hidden Markov Model (HMM). However, other models may be used under the present invention including segment models. Typically, feature vectors can be evaluated against the model, giving a probability that each feature vector was spoken by the same speaker who trained the model. Some models are dependent on what is spoken (so-called text-dependent), other types of models (text-independent) simply evaluate whether any sequence of sounds came from the same speaker who trained the model.

Once the acoustic model has been trained, spectral adder 516 provides the matched test signal to feature extractor 410 which extracts the same type

of features from the matched test signal that were extracted from the matched training signal.

At step 312 of FIG. 3, the extracted features from the matched test signal are applied to acoustic model 418 by a decoder 412. Using acoustic model 418, decoder 412 determines an overall probability that the matched test signal was generated by the same speaker who trained acoustic model 418. This output probability can either be compared to other probabilities generated for other sets of training data or can be compared to a threshold value to determine whether the probability provided by decoder 412 is sufficiently high to identify the speaker as the same person who trained the current model.

Note that in the method of FIG. 3, the model is trained using matched training data that has had its noise matched to the noise in the matched test data. Also note that the matched test data is applied to the model. Under the present invention, such matching is thought to provide a more accurate probability measure for speaker identification.

Step 306 of FIG. 3, which shows the step of adjusting the variances and means of the noise in the training and test signals, represents a step of spectral addition that is performed in order to match the noise in the training signal to the noise in the test signal. Specifically, this step hopes to match the mean strength of each frequency in the noise of the test signal to the mean strength of each

frequency in the noise of the training signal and to match the variance in the strength of each frequency component in these signals.

Under most embodiments of the present invention, the matching is performed by first identifying which signal has the higher mean strength for each frequency component and which signal has the higher variance for each frequency component. The test signal and the training signals are then modified by adding properly adjusted noise segments to each signal so that the mean and variance of each frequency component in the modified signals are equal to the maximum mean and maximum variance found in either signal. Under one embodiment, a cross-condition is applied so that the noise segments that are added to the test signal come from the training signal and the noise segments that are added to the training signal come from the test signal.

As an example, let us say that at frequency F1, the training noise has a mean of 5 and a variance of 2, the testing noise has a mean of 4 and a variance of 3. The following noise will be added to the training signal: test noise at frequency F1 modified such that when added to the training signal, the combined signal will have mean 5 (the greater of the two means) and variance 3 (the greater of the two variances). Thus, the signal to add will have mean 0 and variance 1, since the mean of summed signals is always additive, and the variance of summed linearly independent signals is additive (see Fundamentals of

Applied Probability Theory, Alvin M. Drake, McGraw-Hill Book Company, 1988, p 108). In order to make the test noise segment have these characteristics, the noise segment is shifted and scaled as discussed further below.

Similarly, the noise segment added to the test signal will be a training noise segment that has been scaled and shifted to have a mean of 1 and a variance of zero. When added to the test signal, the noise segment will cause the modified test signal to have mean 5 and variance 3 just like the modified training signal. As will be shown below, this technique of always selecting the signal with the higher mean or higher variance as the signal to match to, eliminates the need for flooring that causes spectral subtraction to be less than ideal.

The means and variances may be adjusted independently by adding two different respective signals to both the test speech signal and training speech signal or at the same time by adding one respective signal to both the test speech signal and the training speech signal. In embodiments where two signals are used, the mean may be adjusted before the variance or after the variance. In addition, the means and variances do not have to both be adjusted, one may be adjusted without adjusting the other. In the discussion below, the embodiment in which two different signals are applied to both the test signal and the training signal is described. In this embodiment, signals to match the variances are first

added to the speech signal and then signals to match the means are added to the speech signals.

The steps for adjusting the variance for a single frequency component are shown in FIG. 7. The method of FIG. 7 begins at step 700 where the variance of the noise in the training signal is determined. To determine the variance of a particular frequency component in the noise of the training signal, the method tracks strength values (i.e. amplitude values or energy values) of this frequency component in different noise segments stored in noise storage 510 of FIG. 5. Methods for determining the variance of such values are well known.

An example of how a frequency component's strength values change over time is shown graph 804 of FIG. 8 where time is shown along horizontal axis 800, and the strength of the frequency component is shown along vertical axis 802. Note that although graph 804 is depicted as a continuous graph, the strength values can be either discrete or continuous under the present invention.

To calculate the complete variance in the noise of the training signal, the strength of the frequency component is measured at each noise frame in the entire training corpus. For example, if the user repeated the identification phrase three times during training, the variance would be determined by looking at all of the noise frames found in the three repetitions of the training phrase.

After the variance of the frequency component in the noise of the training signal has been determined, the method of FIG. 7 continues at step 702 where the variance of the frequency component in the noise of the test signal is determined. The variance of the frequency component in the noise of the test signal is determined using the same techniques discussed above for the training signal.

Once the variances of the frequency component have been determined for the training signal and the test signal, the present invention determines which signal has the greater variance and then adds a noise segment to the other signal to increase the variance of the frequency component in the signal that has the lesser variance so that its variance matches the other signal. For example, if the variance of the frequency component in the noise of the training signal were less than the variance of the frequency component in the noise of the test signal, a modified noise segment from the test signal would be added to the training signal so that the variance in the training signal matches the variance in the test signal.

Under one embodiment, the noise segments are not added directly to the signals to change their variance. Instead the mean strength of the frequency component is set to zero across the noise segment and the variance of the noise segment is scaled. These changes limit the size of the strength values that

are added to the test signal or training signal so that the variances in the test signal and training signal match but the mean strength in the two signals is not increased any more than is necessary. The process of selecting a noise segment, setting the mean of the noise segment's frequency component to zero, and scaling the variance of the noise segment's frequency component are shown as steps 704, 706, 708 and 710 in FIG. 7.

First, at step 704, a noise segment is selected from the testing signal to be added to the training signal and from the training signal to be added to the test signal. These noise segments typically include a plurality of frames of the training signal or testing signal and can be taken from noise storage 510 of FIG. 5. Specifically, the strength values for the current frequency component are retrieved.

An example of how the frequency component's strength for such a selected noise segment changes over time is shown as graph 904 in FIG. 9. In FIG. 9, time is shown along horizontal axis 900 and the strength of the frequency component is shown along vertical axis 902. Although graph 904 shows the frequency component as continuous, the frequency component may be continuous or discrete under the present invention.

After the noise segment has been selected, the mean of the strength of the frequency component

in the noise segment is determined at step 706. In FIG. 9, the mean is shown as horizontal line 906.

In step 708 of FIG. 7, the mean determined in step 706 is subtracted from each of the strength values of the frequency component across the entire noise segment. For an embodiment where the strength values are continuous, this involves subtracting line 906 of FIG. 9 from graph 904 of FIG. 9. This subtraction results in a set of modified strength values for the frequency component of the noise segment. A graph 1004 of such modified strength values is shown in FIG. 10 where time is shown along horizontal axis 1000 and strength is shown along vertical axis 1002.

The mean strength of the frequency component is subtracted from the frequency component's strength values in order to generate a set of strength values that have zero mean but still maintain the variance found in the original noise segment. Thus, in FIG. 10, the strength of the frequency component continues to vary as it did in the original noise segment, however, its mean has now been adjusted to zero.

In step 710, once the values of the frequency component's strength have been adjusted so that they have zero mean, the values are scaled so that they provide a proper amount of variance. This scaling factor is produced by multiplying each of the strength values by a variance gain factor. The

variance gain factor, G, is determined by the following equation:

$$G = \frac{\left|\sigma^2_{TRAIN} - \sigma^2_{TEST}\right|}{\sigma^2_{NOISE}} \qquad \text{Eq. 1}$$

where G is the variance gain factor, σ^2_{TRAIN} is the variance in the training signal, σ^2_{TEST} is the variance in the test signal, and σ^2_{NOISE} is the variance of the values in the zero-mean noise segment produced at step 708.

The result of multiplying strength values by the gain factor of equation 1 is shown in graph 1104 of FIG. 11 where time is shown along horizontal axis 1100 and strength is shown along vertical axis 1102. The frequency component values has the same general shape as graph 1004 of FIG. 10 but is simply magnified or compressed.

After step 710, the modified frequency component values of the noise segment have zero mean and a variance that is equal to the difference between the variance of the training signal and the variance of the test signal. Thus, the modified values can be thought of as a variance pattern. When added to the signal with the lesser variance the strength values of this variance pattern cause the signal with the lesser variance to have a new variance that matches the signal with the larger variance. For example, if the test signal had a lower variance than the training signal, adding the

variance pattern from the training noise segment to each of a set of equally sized segments in the test signal would generate a test signal with a variance that matches the higher variance of the training signal. The step of adding the variance pattern to the strength values of the test signal or training signal is shown as step 712.

Note that for the signal with the higher variance, the variance gain factor is set to zero. When multiplied by the strength values of the noise segment, this causes the modified noise segment to have a mean of zero and a variance of zero.

Note that because of the subtraction performed in step 708, the test signal or training signal produced after step 712 may have a negative strength for one or more frequency components. For example, FIG. 12 shows strength values for the frequency component of either the test signal or training signal, with time shown along horizontal axis 1200 and strength shown along vertical axis 1202. Since the strength values in FIG. 12 are taken from an actual test signal or training signal, all of the strength values in graph 1204 are positive. However, FIG. 13 shows the result of the addition performed in step 712 where the strength values in segments of the test signal are added to respective strength values of the variance pattern shown in FIG. 11. In FIG. 13, time is shown along horizontal axis 1300 and strength is shown along vertical axis 1302. Graph 1304 of FIG. 13 represents the addition of

graph 1104 of FIG. 11 with graph 1204 of FIG. 12. As shown in FIG. 13, graph 1304 includes negative values for some strengths of the frequency component because the variance pattern included some negative values after the mean was subtracted in step 708.

Since a negative strength (either amplitude or energy) for a frequency component cannot be realized in a real system, the strength values for the frequency component in the test signal and training signal must be increased so all of the values are greater than or equal to zero. In addition, the strength values must be increased uniformly so that the variance is unaffected.

To do this, one embodiment of the present invention searches for the most negative value in the entire signal that had its variance increased. This minimum value is shown as minimum 1306 in FIG. 13. Once the minimum value has been identified, its absolute value is added to each of the strength values for the frequency component across the entire test signal and the entire training signal. This is shown as step 716 in FIG. 7.

FIG. 14 provides a graph 1404 of the signal of FIG. 13 after this addition, showing that the strength for the frequency component now has a minimum of zero. In FIG. 14, time is shown along horizontal axis 1400 and strength is shown along vertical axis 1402. Since the strength value added to each of the strength values is the same, the

variance of the test signal and training signal are unchanged.

Note that the strength value must be added to both the test signal and the training signal regardless of which signal had its variance increased. If this were not done, the mean of one of the signals would increase while the mean of the other signal would remain the same. This would cause the means to become mismatched.

In FIG. 7, the step of adjusting the modified test signal and training signal to avoid having negative values in those signals has been shown as occurring before the means of the two signals have been matched. In other embodiments, this step is performed after the means have matched. One benefit of waiting to adjust the signals for negative values until after the means have been matched is that the step of matching the means may cause the signals to be increased to the point where they do not include any negative values.

After step 716, the variances of the test signal and the training signal are matched and each signal only has positive strength values for each frequency component.

Note that the steps of FIG. 7 are repeated for each desired frequency component in the test signal and training signal. Also note that the variance for some frequency components will be higher in the test signal than in the training signal, while for other frequency components, the variance in the

test signal will be lower than in the training signal. Thus, at some frequencies, a variance of the noise segment will be added to the test signal, while at other frequencies, a variance from the noise segment will be added to the training signal.

Once the variances in the noise of the training signal and test signal have been matched, the means of the strength values in the noise of the two signals are matched. This is shown as step 308 in FIG. 3 and is shown in detail in the flow diagram of FIG. 15. As in the method of FIG. 7, the steps for matching a mean strength shown in FIG. 15 are repeated for each frequency component of interest in the noise of the test signal and training signal. Consistent with the discussion above, the mean strength can either be the mean amplitude or the mean energy, depending on the particular embodiment.

In step 1500 of FIG. 15, the mean strength of the current frequency component in the noise of the training signal is determined. This mean can be calculated by accessing the strength values stored in noise storage 506 for the noise segments of the training signal. In step 1502, the mean strength for the current frequency component in the noise of the test signal is determined by accessing the strength values found in noise storage 508.

In step 1504 of FIG. 15, the difference between the means in the noise of the test signal and the training signal are determined. This involves simply subtracting the mean strength of one signal

from the other and taking the absolute value of the result.

In step 1506, the signal with the lower mean has all of its strength values for the frequency component increased by an amount equal to the difference between the means of the test signal and the training signal. This can be seen by comparing FIGS. 16 and 17. In FIG. 16, graph 1604 shows the strength of a frequency component of the test signal or training signal as a function of time. In FIG. 16, time is shown along horizontal axis 1600 and strength is shown along vertical axis 1602. FIG. 17 shows the same frequency component for the same signal after the difference between the means of the test signal and training signal has been added to the signal of FIG. 16. Thus, graph 1704 of FIG. 17 has the same shape of graph 1604 of FIG. 16 but is simply shifted upward. This upward shift does not change the variance, but simply shifts the mean of the frequency component across the signal. Thus, the variances continue to be matched after the steps of FIG. 15.

Note that for some frequency components, the mean of the frequency component in the test signal is greater than the mean of the frequency component in the training signal while at other frequencies the reverse is true. Thus, at some frequencies, the difference between the means is added to the test signal while at other frequencies

the difference between the means is added to the training signal.

As mentioned above, in alternative embodiments, only one respective noise signal is added to each of the training signal and test signal in order to match both the variance and means of those signals. Thus, one noise signal generated from a training noise segment would be added to the test signal and one noise signal generated from a test noise segment would be added to the training signal. Under one embodiment, the one noise signal to be added to each speech signal is formed by adding the difference between the means to all of the values of the variance pattern of the signal with the lower mean. The resulting mean adjusted variance pattern is then added to its respective signal as described above.

From the above discussion it can be seen that after the steps of FIG. 7 and FIG. 15 have been performed for each frequency component in the test signal and training signal, the mean of each frequency component in the noise of the training and test signals is the same and the variance of the frequency component in the noise of each signal is the same. This means that the noise in the training data and test data are matched.

Multiple training signals can be dealt with in several ways. Two primary ways are discussed here. First, if all the training signals are considered to have been generated in the same noisy

environment, they can be considered to be one training signal for the above description. If they might have come from separate noisy environments, such as would occur if they were recorded at separate times, the above description would simply be extended to multiple signals. The mean and variance of each frequency of the noise of all signals would be appropriately adjusted (through adding noise from the other conditions) to have the maximum mean and variance at each frequency of any of the multiple signals.

Although the present invention has been described with reference to particular embodiments, workers skilled in the art will recognize that changes may be made in form and detail without departing from the spirit and scope of the invention.

WHAT IS CLAIMED IS:

1. A method of speaker recognition that generates a likelihood that the same speaker generated a training signal and a test signal, the method comprising:

generating a matched test signal and a matched training signal by performing steps for each of a plurality of frequency components, the steps comprising:

adding to the strength of the frequency component in one of the test signal or training signal as part of the production of the matched test signal and matched training signal so that the mean strength of the frequency component of noise in the matched test signal matches the mean strength of the frequency component of noise in the matched training signal;

creating a model based on the matched training signal; and

applying the matched test signal to the model to produce the likelihood that a same speaker generated the training signal and the test signal.

2. The method of claim 1 wherein performing steps for each of a plurality of frequency components, further comprises the steps of:

determining the mean strength of the frequency component of noise in the training signal;

determining the mean strength of the frequency component of noise in the test signal, and

subtracting the mean strength of noise in the training signal from the mean strength of noise in the test signal to determine a value to add during the step of adding to the strength of the frequency component in one of the test signal or training signal.

3. The method of claim 1 wherein for each frequency component the step of adding to the strength of the frequency component in one of the test signal or training signal comprises adding to the strength of the frequency component in the test signal.

4. The method of claim 1 wherein for each frequency component the step of adding to the strength of the frequency component in one of the test signal or training signal comprises adding to

the strength of the frequency component in the training signal.

5. The method of claim 1 wherein for some frequency components the step of adding to the strength of the frequency component in one of the test signal or training signal comprises adding to the strength of the frequency component in the training signal and for other frequency components the step of adding to the strength of the frequency component in one of the test signal or training signal comprises adding to the strength of the frequency component in the test signal.

6. The method of claim 1 wherein adding to the strength of the frequency component in one of the test signal or training signal does not change the variances of the frequency component in the test signal and the training signal.

7. The method of claim 1 wherein generating a matched test signal and a matched training signal by performing steps for each of a plurality of frequency components further comprises:

determining the variance of the frequency component of noise in the training signal;

determining the variance of the frequency component of noise in the test signal; and

increasing the variance of the frequency component in one of the test signal or the training signal as part of the production of the matched test signal and the matched training signal so that the variance of the frequency component in the noise of the matched training signal matches the variance of the frequency component in the noise of the matched test signal.

8. The method of claim 7 wherein increasing the variance of the frequency component in one of the test signal or the training signal comprises:

deriving a variance pattern from a noise segment taken from one of the test signal or training signal; and

adding the variance pattern to all segments of one of the test signal or the training signal.

9. The method of claim 8 wherein deriving the variance pattern of the noise segment comprises:

determining the mean of the noise segment;

subtracting the mean of the noise segment from the noise segment to produce a zero-mean noise segment; and

multiplying the zero-mean noise segment by a gain factor to produce the variance pattern.

10. The method of claim 9 wherein adding the variance pattern to all segments of one of the test signal or training signal further comprises:

after adding the variance pattern determining the most negative value for the frequency component in one of the test signal or the training signal; and

adding a value equal to the magnitude of the most negative value to the frequency component of both the test signal and the training signal.

11. The method of claim 9 wherein adding the variance pattern to all segments of one of the test signal or training signal further comprises:

after adding the variance pattern and adding to the strength of the frequency component in one of the test signal or training signal, determining the most negative value for the frequency component in one of the test signal or the training signal; and

adding a value equal to the magnitude of the most negative value to the frequency component of both the test signal and the training signal.

12. The method of claim 8 wherein adding to the strength of the frequency component in one of the test signal or training signal comprises adding to the strength of the variance pattern before adding the variance pattern to all segments of one of the test signal or training signal.

13. The method of claim 8 wherein deriving a variance pattern comprises deriving a variance pattern from a noise segment taken from the test signal and wherein adding the variance pattern comprises adding the variance pattern to all segments of the training signal.

14. The method of claim 8 wherein deriving a variance pattern comprises deriving a variance pattern from a noise segment taken from the training signal and wherein adding the variance pattern comprises adding the variance pattern to all segments of the test signal.

15. A method of identifying a speaker comprising:

receiving a training speech signal;

receiving a test speech signal;

for each of a plurality of frequency components adding to the variance of the frequency component in one of the training speech signal or test speech signal so that the variance of the

frequency component is matched in a matched training speech signal and a matched test speech signal;

generating a model from the matched training speech signal; and

comparing the matched test speech signal to the model to identify the speaker.

16. The method of claim 15 wherein adding to the variance of the frequency component comprises:

identifying a series of strength values for the frequency component in a segment of noise taken from one of the training speech signal or the test speech signal;

finding the mean of the series of strength values;

subtracting the mean from each strength value in the series of strength values to generate zero-mean strength values;

multiplying the zero-mean strength values by a gain factor to produce a variance pattern; and

adding the variance pattern to each segment of one of the training speech signal or the test speech signal.

17. The method of claim 16 wherein after adding the variance pattern the method further comprises:

determining the most negative value for the strength of the frequency component in the one of the training speech signal or test speech signal to which the variance pattern was added; and

adding the absolute value of the most negative value to the strength of the frequency component over the entire training speech signal and the entire test speech signal.

18. The method of claim 15 further comprising for each of a plurality of frequency components adding to the frequency component of one of the test speech signal or the training speech signal so that the mean strength of the frequency component in the test speech signal is matched to the mean strength of the frequency component in the training speech signal.

19. The method of claim 16 further comprising before adding the variance pattern to each segment of one of the training speech signal or test speech signal:

adding a same value to each strength value of the variance pattern so that the mean strength of the frequency component in the test speech signal is matched to the mean strength of the frequency component in the training

speech signal when the variance pattern is added to each segment of one of the training speech signal or the test speech signal.

20. The method of claim 15 further comprising:

receiving a second training speech signal; and

wherein adding to the variance of a frequency component in one of the training speech signal or test speech signal comprises:

identifying the largest variance in the test speech signal, the training speech signal and the second training speech signal; and

adjusting the variance of the test speech signal, the training speech signal and the second training speech signal to match the largest identified variance.

21. A computer-readable medium having computer-executable instructions for performing speaker recognition, the instructions performing steps comprising:

receiving a training speech signal;

receiving a test speech signal;

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adding to the strength of at least one frequency component across the entirety of one of the training speech signal or test speech signal in the production of a matched training speech signal and a matched test speech signal such that the mean strength of the frequency component in noise in the matched training speech signal is the same as the mean strength of the frequency component in noise in the matched test speech signal;

generating a model from the matched training speech signal; and

comparing the matched test speech signal to the model to identify a speaker.

22. The computer-readable medium of claim 21 wherein adding to the strength of a frequency component comprises:

determining the mean strength of the frequency component in noise in the training speech signal;

determining the mean strength of the frequency component in noise in the test speech signal;

determining the difference between the mean strength in noise in the training

speech signal and the mean strength in noise in the test speech signal;

adding the difference to the strength of the frequency component in one of the training speech signal or test speech signal.

23. The computer-readable medium of claim 21 wherein the computer-executable instructions perform further steps comprising:

selectively adding to the strength of the frequency component in one of the training speech signal or test speech signal in further production of the matched training speech signal and the matched test speech signal such that the variance of the strength of the frequency component of noise in the matched training speech signal is equal to the variance of the strength of the frequency component of noise in the matched test speech signal.

24. The computer-readable medium of claim 23 wherein selectively adding to the strength of the frequency component comprises:

selecting a noise segment from one of the training speech signal or the test speech signal;

identifying strength values of the frequency component in the noise segment;

determining the mean of the strength values;

subtracting the mean of the strength values from the strength values to produce a sequence of mean adjusted strength values;

multiplying the mean adjusted strength values by a gain factor to produce gain adjusted strength values;

adding the gain adjusted strength values to respective strength values of the frequency component in each of a plurality of segments that together constitute one of the training speech signal or test speech signal.

25. The computer-readable medium of claim 24 wherein adding to the strength of at least one frequency component across the entirety of one of the training speech signal or test speech signal comprises adding the same value to all of the gain adjusted strength values before adding the gain adjusted strength values to the respective strength values.

26. The computer-readable medium of claim 25 wherein selectively adding to the strength of the frequency component further comprises:

identifying the most negative value produced by adding the gain adjusted strength values to the respective strength values of the frequency component in each of a plurality of segments that constitute one of the training speech signal and test speech signal; and

adding a value equal to the absolute magnitude of the most negative value to each strength value of the frequency component in both the training speech signal and the test speech signal.

27. The computer-readable medium of claim 24 wherein the computer-executable instructions perform further steps comprising:

determining the variance of strength values of the frequency component in the test speech signal;

determining the variance of strength values of the frequency component in the training speech signal;

determining the variance of the strength values of the frequency component in the noise segment; and

determining the gain factor by subtracting the variance in the test speech signal from the variance in the training speech signal and dividing the difference by the variance in the noise segment.

28. The computer-readable medium of claim 24 wherein selectively adding to the strength of the frequency component further comprises:

identifying the most negative value produced by adding the gain adjusted strength values to the respective strength values of the frequency component in each of a plurality of segments that constitute one of the training speech signal and test speech signal; and

adding a value equal to the absolute magnitude of the most negative value to each strength value of the frequency component in both the training speech signal and the test speech signal.

29. An apparatus for recognizing a speaker based on a test signal and a training signal, the apparatus comprising:

a spectral addition component for adding to the strength of a frequency component

in at least one of the test signal or training signal to produce a matched test signal and a matched training signal such that the mean and variance of the strength of the frequency component in noise in the matched training signal are respectively equal to the mean and variance of the strength of the frequency component in noise in the matched test signal;

a trainer for training a speaker recognition model based on the matched training signal; and

a decoder capable of generating a probability for an identity of a speaker based on the model and the matched test signal.

Docket No. 95E4909

METHOD AND APPARATUS USING SPECTRAL ADDITION FOR SPEAKER RECOGNITION

ABSTRACT OF THE DISCLOSURE

A method and apparatus for speaker recognition is provided that matches the noise in training data to noise in testing data using spectral addition. Under spectral addition, the mean and variance for a plurality of frequency components are adjusted in the training data and the test data so that each mean and variance is matched in a resulting matched training signal and matched test signal. The adjustments made to the training data and test data add to the mean and variance of the training data and test data instead of subtracting from the mean and variance.

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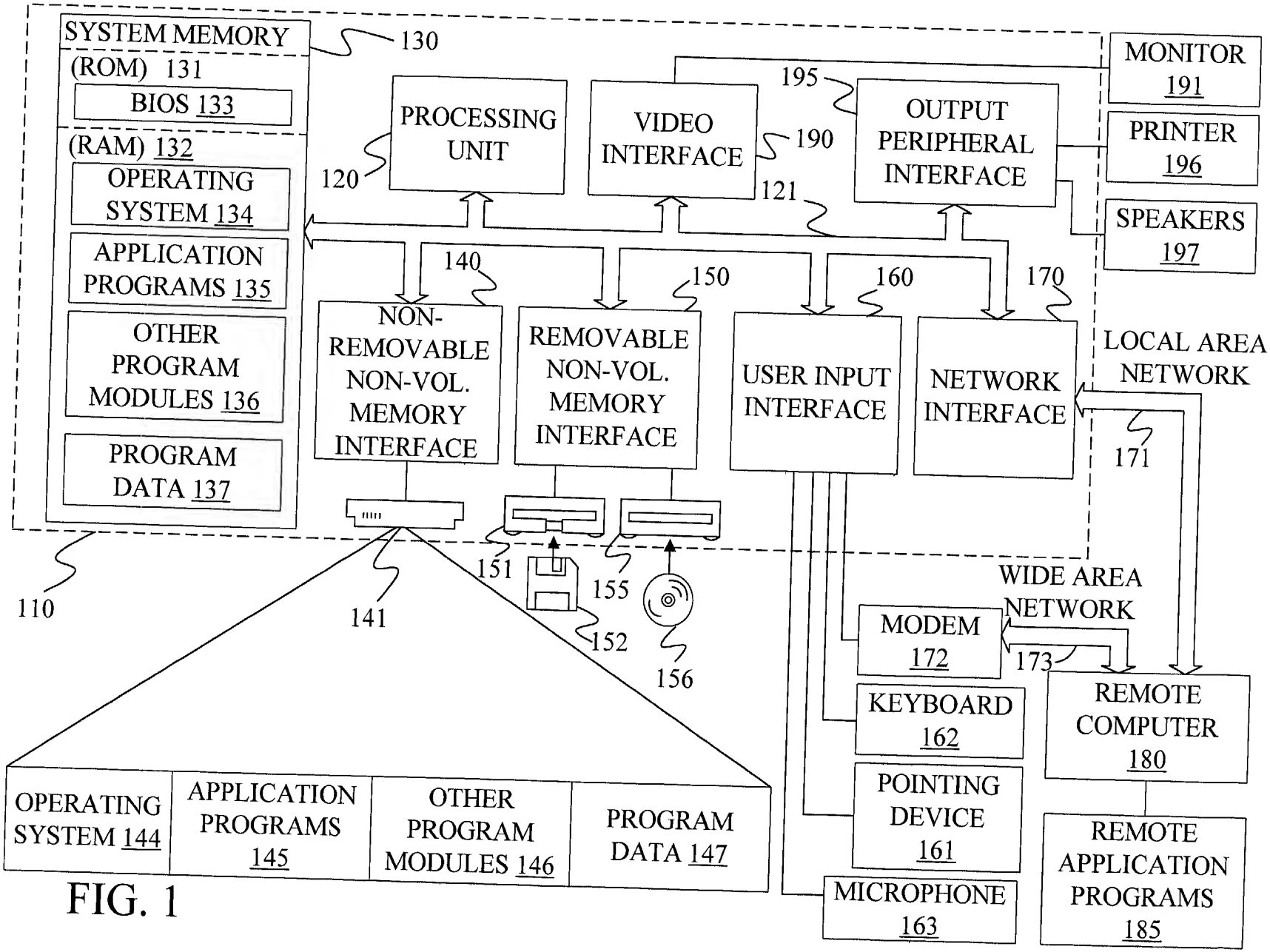

FIG. 1

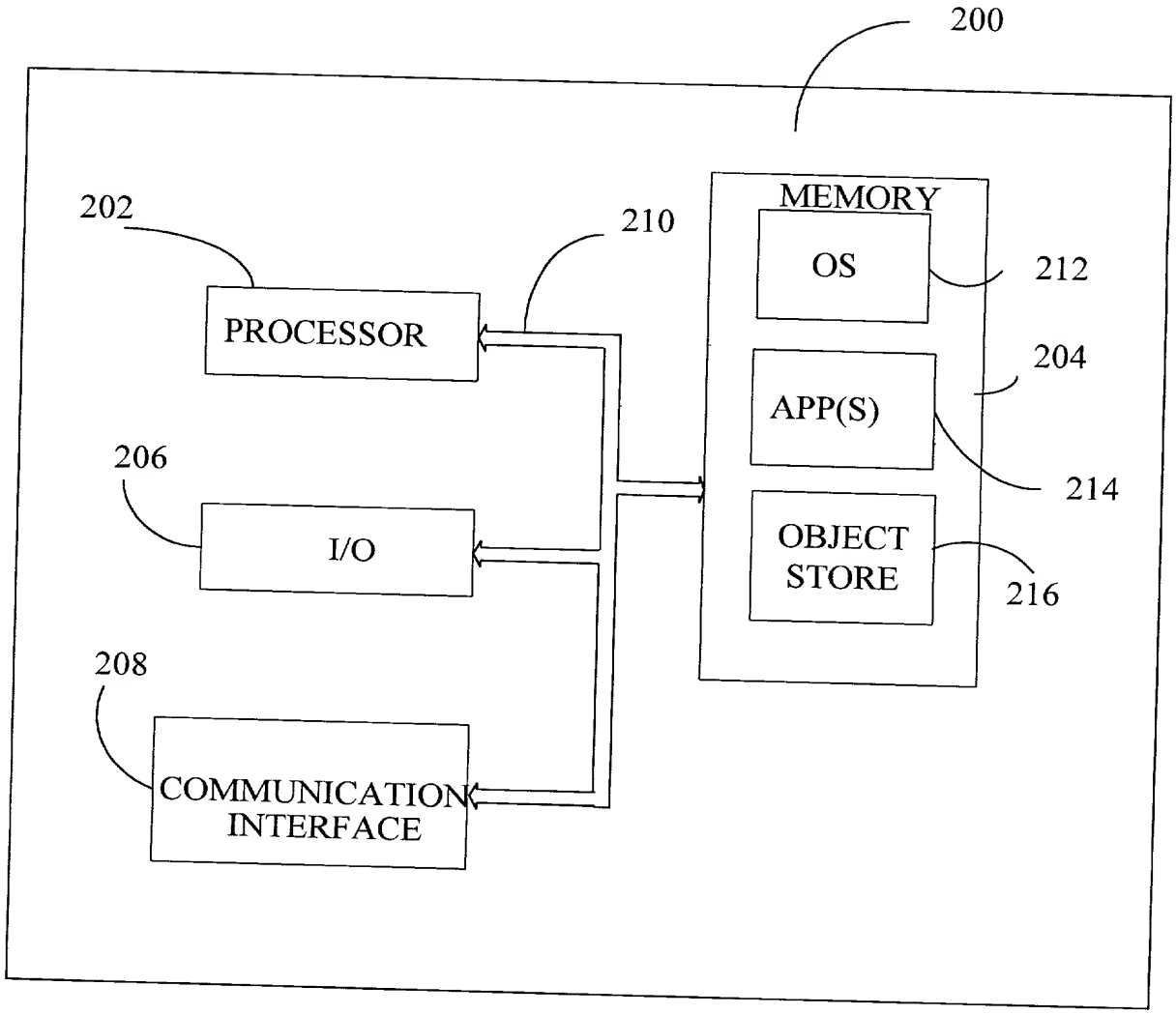
200
202
210
MEMORY
OS
212
PROCESSOR
204
APP(S)
214
206
I/O
OBJECT STORE
216
208
COMMUNICATION INTERFACE

FIG. 2

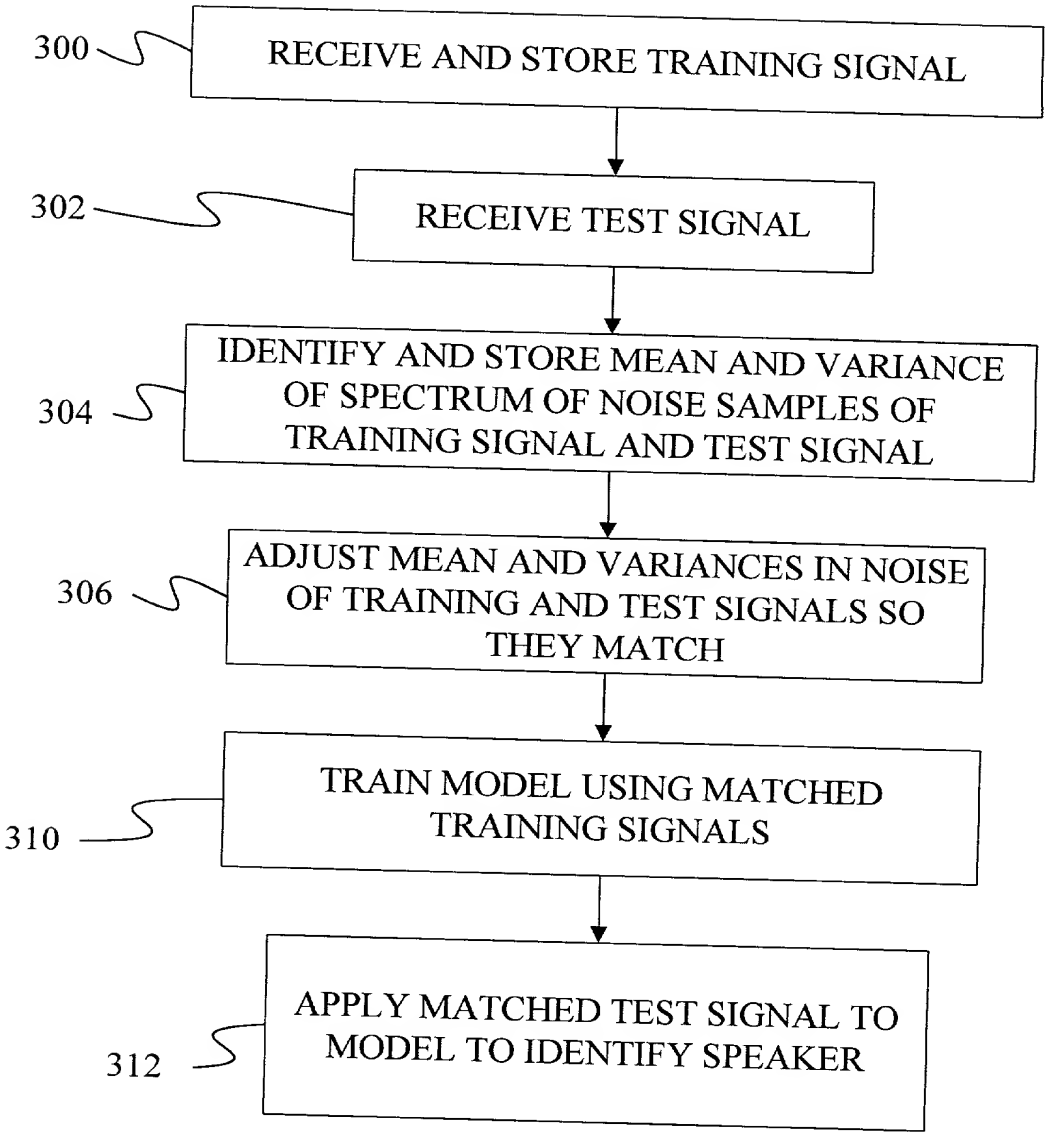
300
RECEIVE AND STORE TRAINING SIGNAL
302
RECEIVE TEST SIGNAL
304
IDENTIFY AND STORE MEAN AND VARIANCE OF SPECTRUM OF NOISE SAMPLES OF TRAINING SIGNAL AND TEST SIGNAL
306
ADJUST MEAN AND VARIANCES IN NOISE OF TRAINING AND TEST SIGNALS SO THEY MATCH
310
TRAIN MODEL USING MATCHED TRAINING SIGNALS
312
APPLY MATCHED TEST SIGNAL TO MODEL TO IDENTIFY SPEAKER

FIG. 3

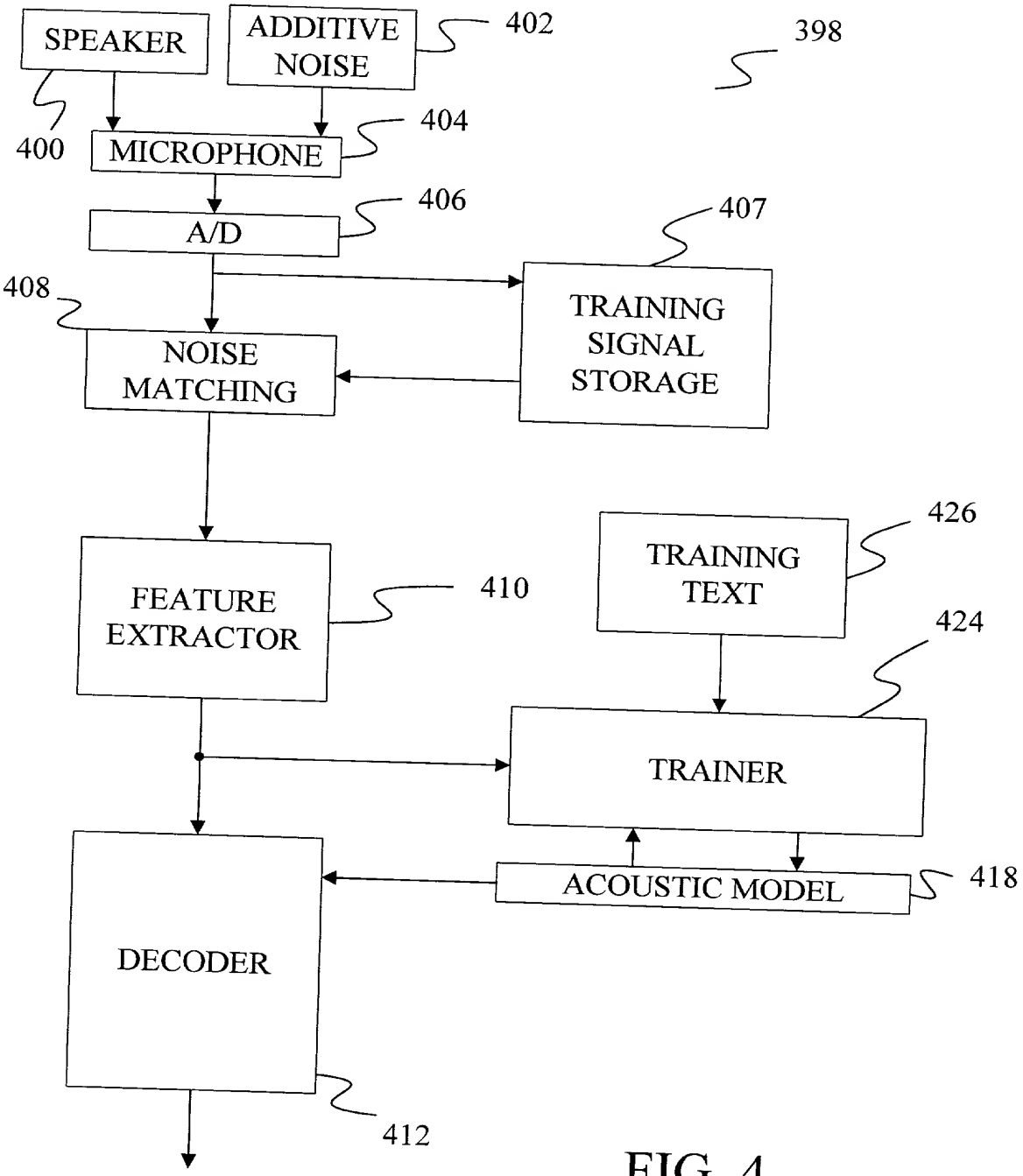
SPEAKER
ADDITIVE NOISE
402
398
400
MICROPHONE
404
A/D
406
407
408
NOISE MATCHING
TRAINING SIGNAL STORAGE
FEATURE EXTRACTOR
410
TRAINING TEXT
426
424
TRAINER
ACOUSTIC MODEL
418
DECODER
412

FIG. 4

0001010 "EP 0558960

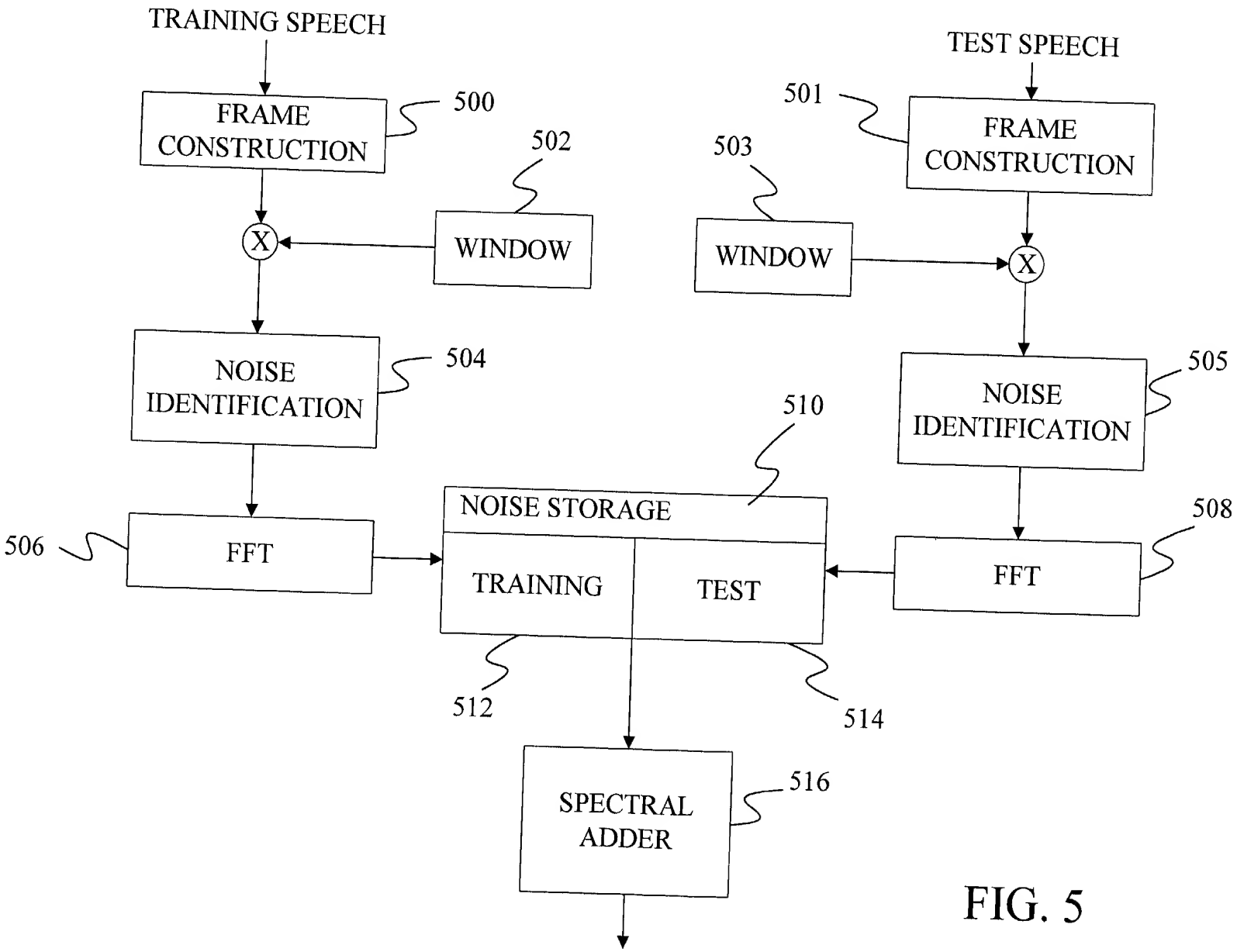

FIG. 5

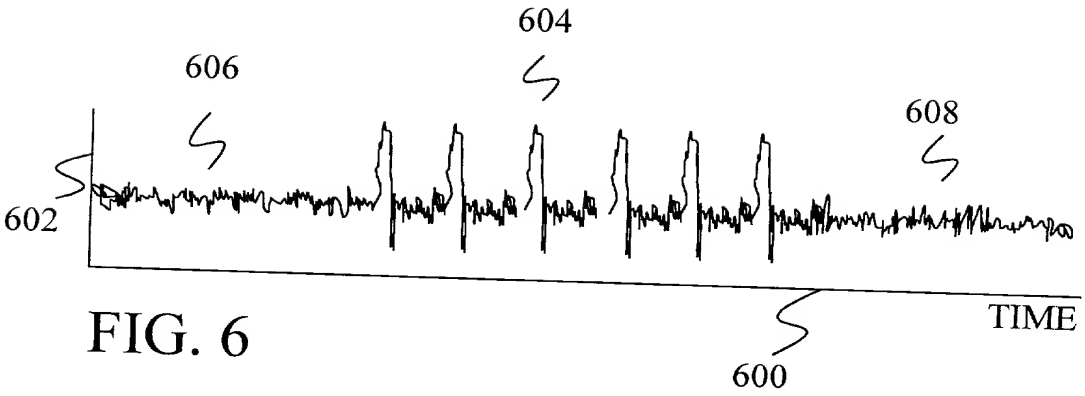

FIG. 6

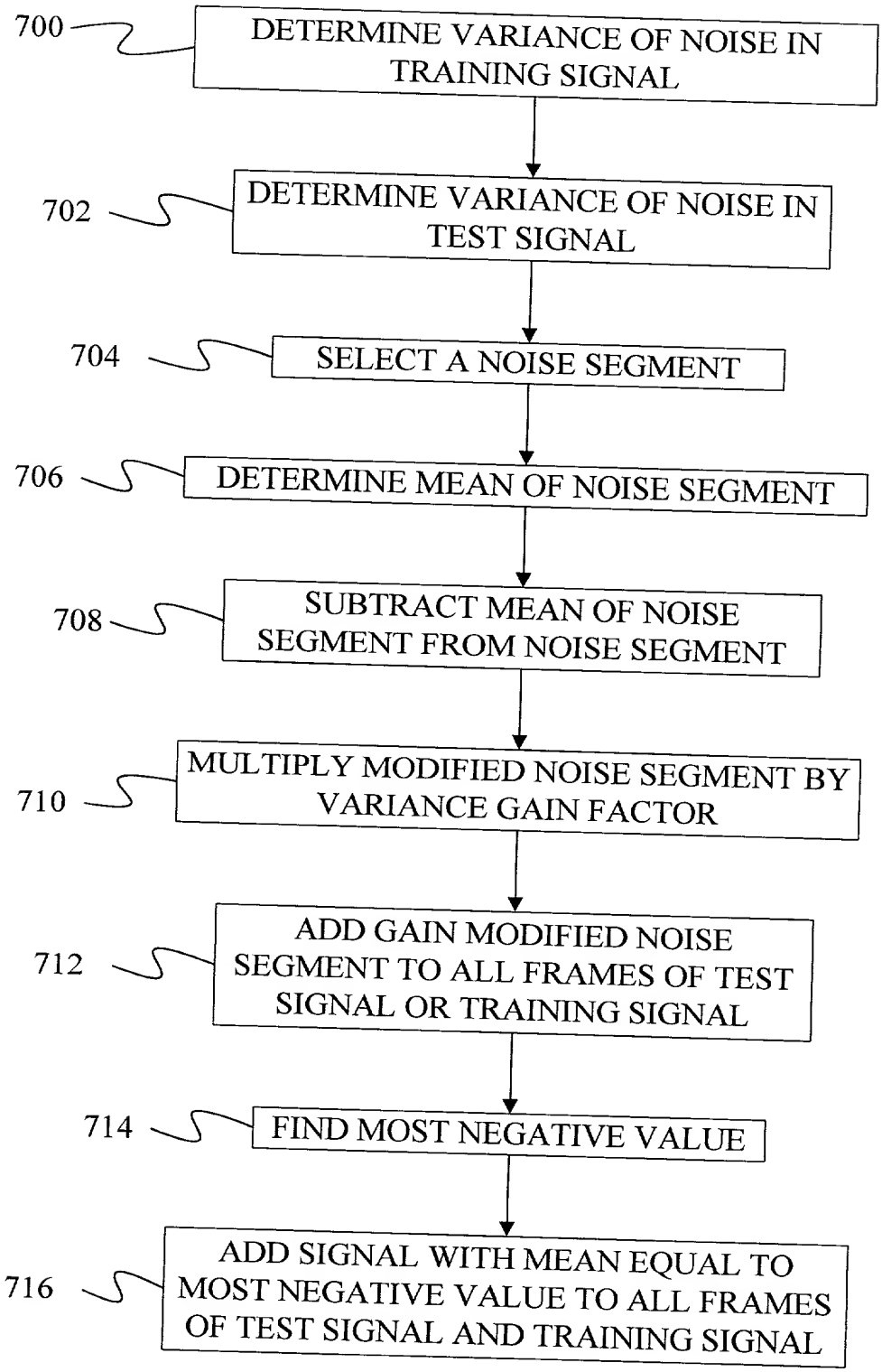
700
DETERMINE VARIANCE OF NOISE IN TRAINING SIGNAL
702
DETERMINE VARIANCE OF NOISE IN TEST SIGNAL
704
SELECT A NOISE SEGMENT
706
DETERMINE MEAN OF NOISE SEGMENT
708
SUBTRACT MEAN OF NOISE SEGMENT FROM NOISE SEGMENT
710
MULTIPLY MODIFIED NOISE SEGMENT BY VARIANCE GAIN FACTOR
712
ADD GAIN MODIFIED NOISE SEGMENT TO ALL FRAMES OF TEST SIGNAL OR TRAINING SIGNAL
714
FIND MOST NEGATIVE VALUE
716
ADD SIGNAL WITH MEAN EQUAL TO MOST NEGATIVE VALUE TO ALL FRAMES OF TEST SIGNAL AND TRAINING SIGNAL

FIG. 7

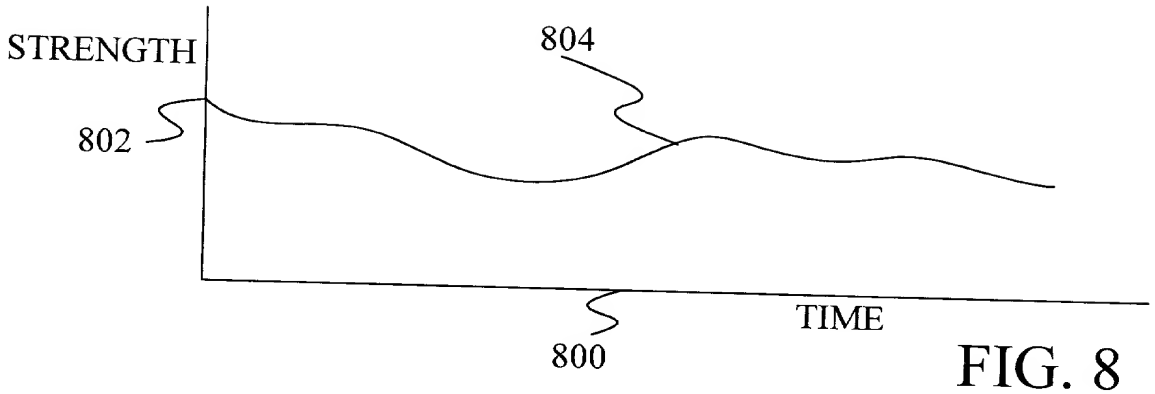

FIG. 8

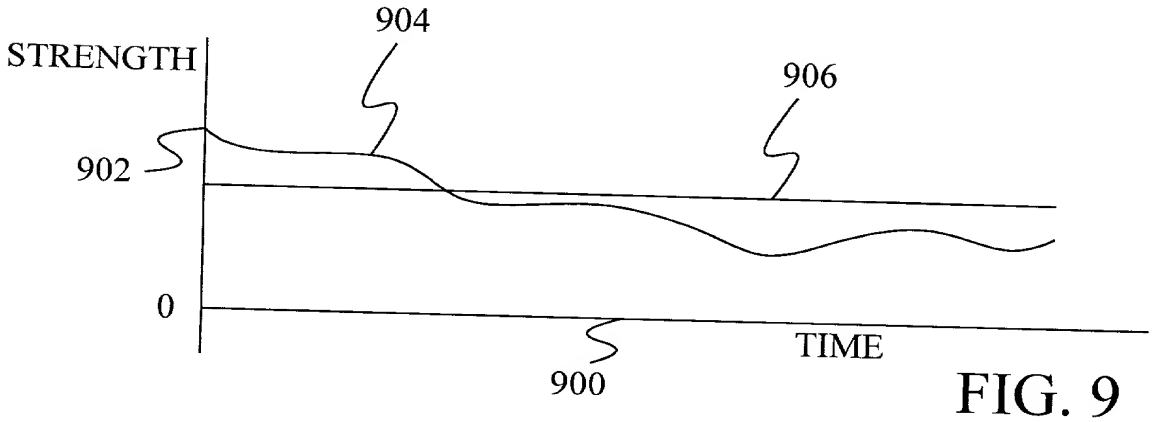

FIG. 9

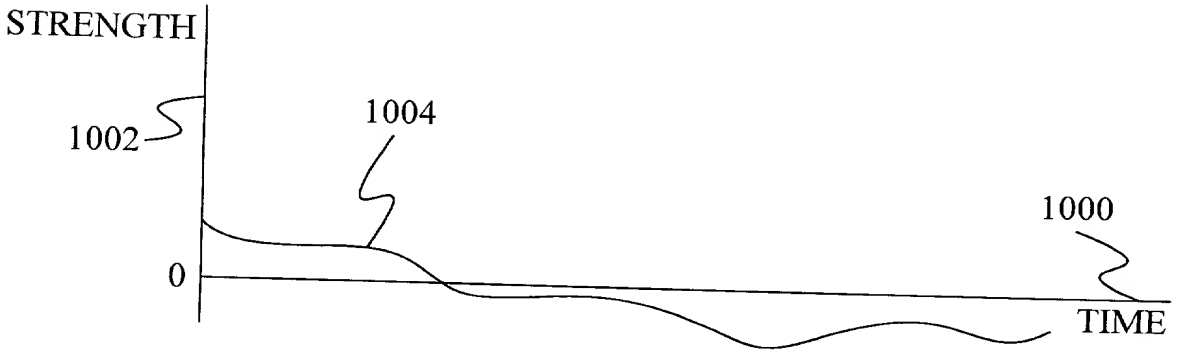

FIG. 10

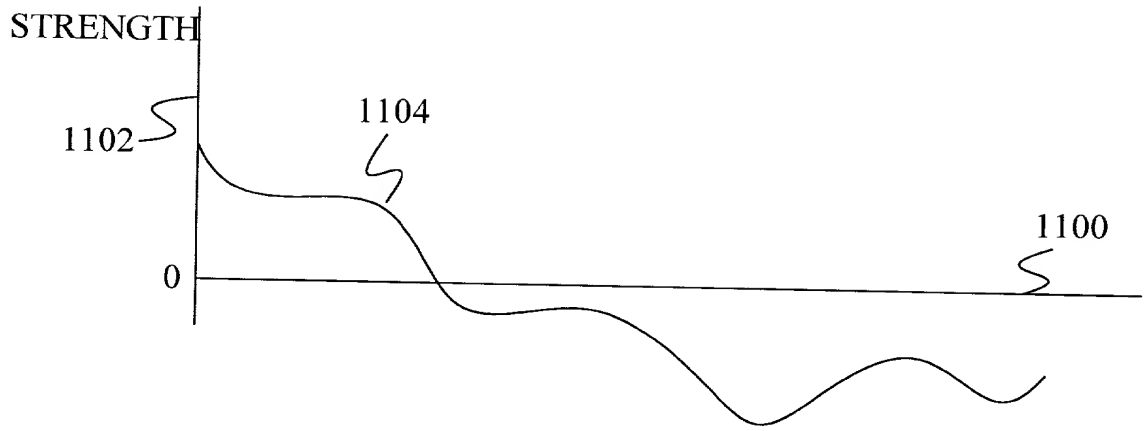

FIG. 11

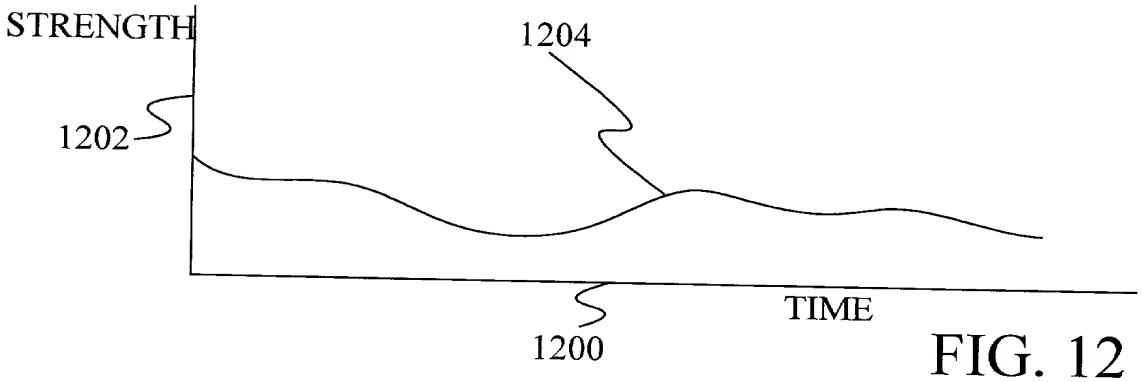

FIG. 12

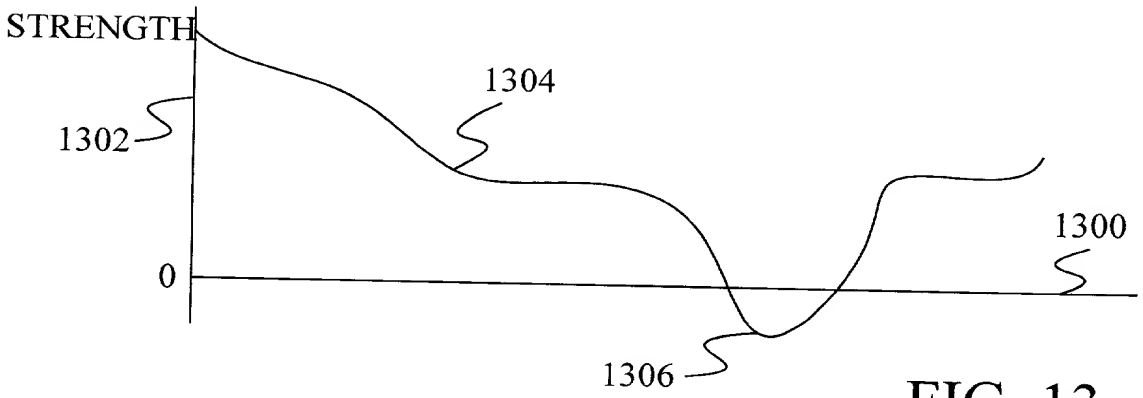

FIG. 13

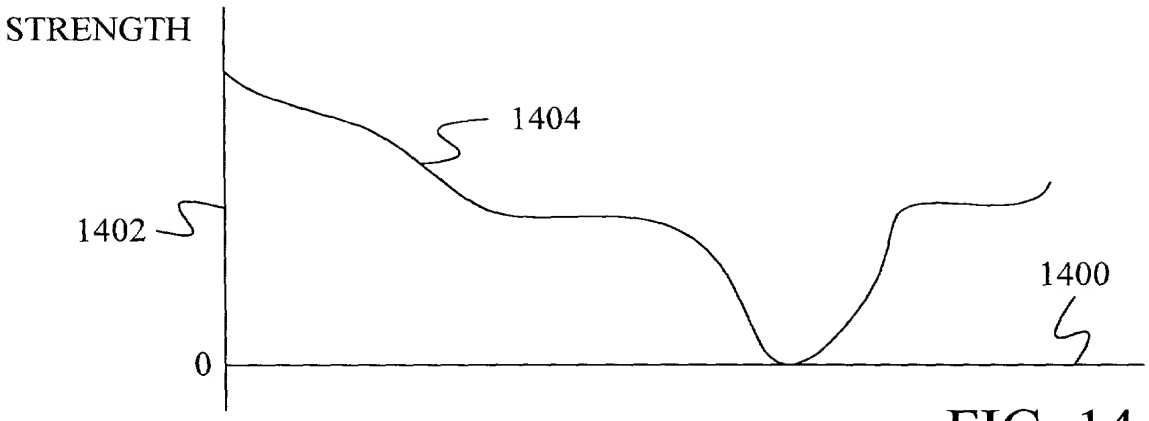
STRENGTH
1404
1402
1400
0

FIG. 14

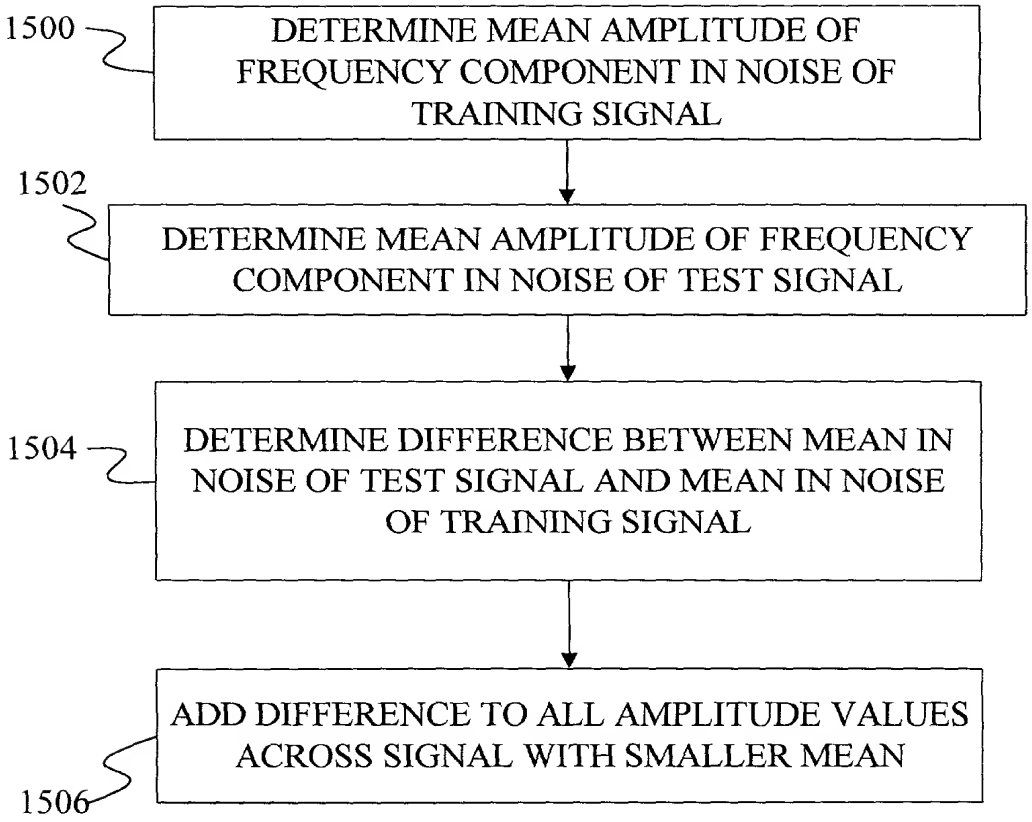
1500
DETERMINE MEAN AMPLITUDE OF FREQUENCY COMPONENT IN NOISE OF TRAINING SIGNAL
1502
DETERMINE MEAN AMPLITUDE OF FREQUENCY COMPONENT IN NOISE OF TEST SIGNAL
1504
DETERMINE DIFFERENCE BETWEEN MEAN IN NOISE OF TEST SIGNAL AND MEAN IN NOISE OF TRAINING SIGNAL
ADD DIFFERENCE TO ALL AMPLITUDE VALUES ACROSS SIGNAL WITH SMALLER MEAN
1506

FIG. 15

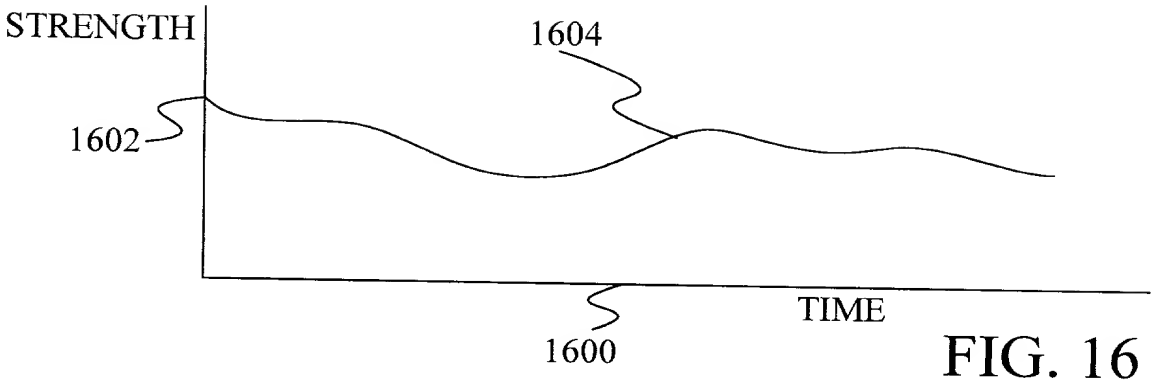

FIG. 16

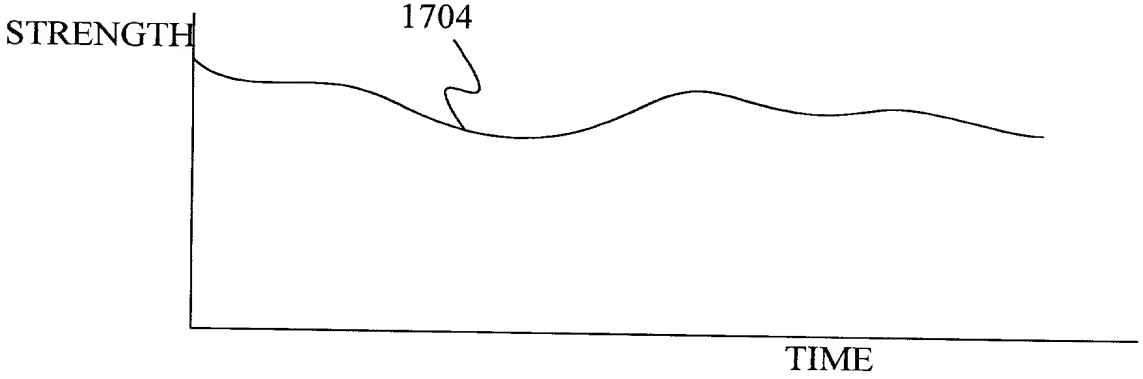

FIG. 17

COMBINED DECLARATION AND POWER OF ATTORNEY IN ORIGINAL APPLICATION	Attorney Docket No. M61.12-0316

SPECIFICATION AND INVENTORSHIP IDENTIFICATION

As a below named inventor, I declare that:

My residence, post office address and citizenship are as stated below next to my name.

I believe I am the original, first and sole inventor of the subject matter which is claimed, and for which a patent is sought, on the invention entitled METHOD AND APPARATUS USING SPECTRAL ADDITION FOR SPEAKER RECOGNITION the specification of which,

(check one) _X_ is attached hereto.
___ was filed on _____ as Appln. No. _____.
___ and was amended on _____.
___ was described and claimed in PCT International Application No. _____ filed on _____ and as amended under PCT Article 19 on _____.

ACKNOWLEDGEMENT OF REVIEW OF PAPERS AND DUTY OF CANDOR

I have reviewed and understand the contents of the above identified specification, including the claims, as amended by any amendment referred to above. I acknowledge the duty to disclose information which is known to me to be material to the patentability of this application in accordance with 37 C.F.R. § 1.56.

PRIORITY CLAIM (35 U.S.C. § 119)

Prior Foreign Application(s)

I claim foreign priority benefits under 35 U.S.C. § 119(a-d) of any foreign application(s) for patent or inventor's certificate listed below and have also identified below any foreign application for patent or inventor's certificate having a filing date before that of the application on which priority is claimed:

Number	Country	Day/Month/Year Filed	Priority Claimed
_____	_____	_____	Yes___ No
_____	_____	_____	Yes___ No

Prior Provisional Application(s)

I hereby claim the benefit under 35 U.S.C. §119(e) of any United States Provisional Application(s) listed below:

Number	Day/Month/Year Filed
_____	_____
_____	

PRIORITY CLAIM (35 U.S.C. § 120)

I claim the benefit under 35 U.S.C. § 120 of any United States application(s) listed below. Insofar as the subject matter of each of the claims of this application is not disclosed in the prior United States application in the manner provided by the first paragraph of 35 U.S.C. § 112, I acknowledge the duty to disclose to the Patent Office all information known to me to be material to patentability as defined in 37 C.F.R. § 1.56 which became available between the filing date of the prior application and the national or PCT international filing date of this application:

Appln. Ser. No.	U.S. Serial No. (if any under PCT)	Filing Date	Status

DECLARATION

I declare that all statements made herein that are of my own knowledge are true and that all statements that are made on information and belief are believed to be true; and further that these statements were made with the knowledge that willful false statements and the like so made are punishable by fine or imprisonment, or both, under 18 U.S.C. § 1001 and that such willful false statements may jeopardize the validity of the application or any patent issued thereon.

POWER OF ATTORNEY

I appoint the following attorneys and agents to prosecute the patent application identified above and to transact all business in the Patent and Trademark Office connected therewith, including full power of association, substitution and revocation: Judson K. Champlin, Reg. No. 34,797; Joseph R. Kelly, Reg. No. 34,847; Nickolas E. Westman, Reg. No. 20,147; Steven M. Koehler, Reg. No. 36,188; David D. Brush, Reg. No. 34,557; John D. Veldhuis-Kroeze, Reg. No. 38,354; Deirdre Megley Kvale, Reg. No. 35,612; Theodore M. Magee, Reg. No. 39,758; Peter S. Dardi, Reg. No. 39,650; Christopher R. Christenson, Reg. No. 42,413; John A. Wiberg, Reg. No. 44,401; Brian D. Kaul, Reg. No. 41,885; Robert M. Angus, Reg. No. 24,383; Katie E. Sako, Reg. No. 32,628; and Daniel D. Crouse, Reg. No. 32,022.

I ratify all prior actions taken by Westman, Champlin & Kelly, P.A. or the attorneys and agents mentioned above in connection with the prosecution of the above-mentioned patent application.

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